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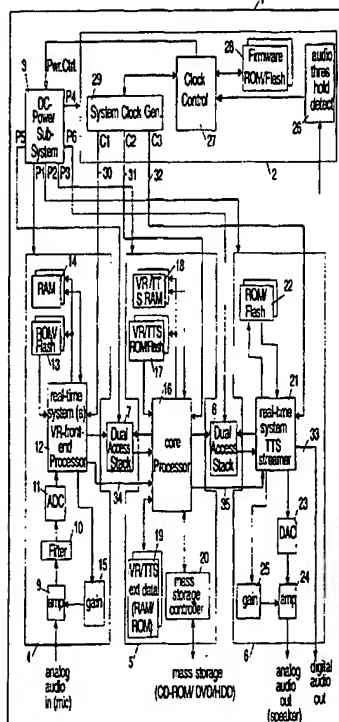
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(54) Title: METHOD AND PROCESSOR SYSTEM FOR PROCESSING OF AN AUDIO SIGNAL



(57) **Abstract:** In a processor system (1) for audio processing, such as voice recognition and text-to-speech, a dedicated front-end processor (12), a core processor (16) and a dedicated back-end processor (21) are provided which are coupled by dual access stack (7) and (8), respectively. When an audio signal is inputted core processor (16) is involved only when a certain amount of data is present in the dual access stack (7). Likewise the back-end processor (21) is involved only when a certain amount of data is present in the dual access stack (8). This way the overall processing power required by the processing task is minimised as well as the power consumption of the processor system (1).

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ABSTRACT:

CHG DATE=20021101 STATUS=N>In a processor system (1) for audio processing, such as voice recognition and text-to-speech, a dedicated front-end processor (12), a core processor (16) and a dedicated back-end processor (21) are provided which are coupled by dual access stack (7) and (8), respectively. When an analog audio signal is inputted core processor (16) is invoked only when a certain amount of data is present in the dual access stack (7). Likewise the back-end processor (21) is invoked only when a certain amount of data is present in the dual access stack (8). This way the overall processing power required by the processing task is minimised as well as the power consumption of the processor system (1).

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